II. CLAIM AMENDMENTS

1. (Original) A method of processing subband signals for cancelling an undesired effect on a signal, the method comprising steps of:

   analysing a primary signal, which has a signal affected by an undesired signal, and a reference signal corresponding to the undesired signal, to produce frequency domain primary signals and frequency domain reference signals in a plurality of subbands;

   processing the frequency domain primary signal and the frequency domain reference signal using an adaptive filter in each subband, comprising operating on at least the frequency domain reference signal to improve the convergence of the adaptive filter in each subband; and

   synthesizing the outputs of the adaptive processing blocks to output a time domain signal in which the effect of the undesired signal has been cancelled.

2. (Currently Amended) A method of claim 1, wherein the operating step includes the step of whitening the input of the adaptive filter by spectral emphasis in the frequency domain.

3.   (Original) A method of claim 1, wherein the operating step includes the step of whitening the input of the adaptive filter by adding noise.

4.   (Original) A method of claim 1, wherein the operating step includes the step of whitening the input of the adaptive filter by decimating the frequency domain primary signal and frequency domain reference signal by a factor (DEC) which is less than or equal to an oversampling factor (OS).

5.   (Original) A method of claim 1, wherein the operating step includes the step of implementing the affine projection algorithm.

6.   (Currently Amended) A method of claim 1, wherein the operating step includes the step of implementing, in each subband, the least mean square algorithm, the recursive least squares algorithm, or a combination thereof.

7.   (Original) A method of claim 1, wherein the operating step includes the step of employing a double-talk detector to control the adaptation process of the adaptive filter.

8.   (Original) A method of claim 1, wherein the operating step includes the step of controlling the adaptation step size of the adaptive filter.

9.   (Original) A method of claim 1, wherein the processing step includes the step of performing non-adaptive noise reduction for eliminating uncorrelated noise.

10.   (Original) A method of claim 1, wherein the processing step includes the step of performing a cross talk resistant adaptive processing using two adaptive filters in each subband.

11.   (Original) A method of claim 1, wherein the undesirable signal is an echo signal or a noise signal.

12.   (Original) A system for processing subband signals for cancelling an undesired effect on a signal, the system comprising:

   an analysis filterbank for analysing a primary signal, which has a signal affected by an undesired signal, and a reference signal corresponding to the undesired signal, to produce frequency domain primary signals and frequency domain reference signals in a plurality of subbands;

a processing module for processing the frequency domain primary signals and the frequency domain reference signals, including an adaptive filter module in each subband, and a module for operating on at least the frequency domain reference signal to improve the convergence of each adaptive filter; and

a synthesis filterbank for synthesizing the outputs of the processing module to output a time domain signal in which the effect of the undesired signal has been cancelled.

13. (Currently Amended) The system according to claim 12, wherein the module whitens the input of the adaptive filter by spectral emphasis in the frequency domain.

14. (Original) The system according to claim 12, wherein the module whitens the input of the adaptive filter by adding noise.

15. (Original) The system according to claim 12, wherein the module whitens the input of the adaptive filter in each subband by decimating the input of the adaptive filter.

16. (Original) The system according to claim 12, wherein the module implements affine projection algorithm in each subband.

17.   (Currently Amended) The system according to claim 12, wherein the module implements the least mean square algorithm, the recursive least squares algorithm, or a combination thereof in each subband.

18.   (Original) The system according to claim 12, wherein the module includes a double-talk detector to control the adaptation process of the adaptive filter.

19.   (Original) The system according to claim 12, wherein the module controls the adaptation step size of each adaptive filter.

20.   (Original) The system according to claim 12, further comprising a non-adaptive noise reduction module for eliminating uncorrelated noise.

21.   (Original) The system according to claim 12, wherein the processing module includes a cross talk resistant adaptive processing module having a pair of adaptive filters in each subband, which process the frequency domain primary signal and frequency domain reference signal.

22.   (Original) The system according to claim 12, wherein the undesirable signal is an echo signal or a noise signal.

23. (New) A method according to claim 1, wherein the analysing step produces oversampled subband signals.

24. (New) A method according to claim 23, wherein the analysing step performs a weighted overlap-added (WOLA) analysis, and the synthesizing step performs a WOLA synthesis.

25. (New) A method according to claim 24, wherein the analysing step includes the steps of:

   performing WOLA analysis on the primary signal in a time
      domain to provide the frequency domain primary
      signals; and

   performing WOLA analysis on the reference signal in the
      time domain to provide the frequency domain reference
      signals.

26. (New) A method according to claim 25, wherein the operating step includes the step of adjusting coefficients of the adaptive filter based on at least the frequency domain reference signal.

27. (New) A method according to claim 25, wherein the adjusting step adjusts the coefficients of the adaptive

filter based on the frequency domain primary signal and the frequency domain reference signal.

28. (New) A system according to claim 12, wherein the analysis filterbank includes an oversampled filterbank having the analysis filerbank for producing oversampled subband signals and the synthesis filterbank for synthesizing the outputs of the processing module.

29. (New) A system according to claim 28, wherein the analysis filterbank includes a weighted overlap-added (WOLA) analysis filterbank, and the synthesis filterbank includes a WOLA synthesis filterbank

30. (New) A system according to claim 29, wherein the analysis filterbank includes:

   a first WOLA analysis filterbank for providing the frequency domain primary signals; and

   a second WOLA analysis filterbank for providing the frequency domain reference signals.

31. (New) A system according to claim 30, wherein the operating module adjusts coefficients of the adaptive filter based on at least the frequency domain reference signal.

32.   (New) A system according to claim 30, wherein the operating module adjusts the coefficients of the adaptive filter based on the frequency domain primary signal and the frequency domain reference signal.